

## **METHOD AND APPARATUS FOR FORMING MULTIPLE BEAMS**

### **Cross-Reference to Related Application**

This application claims the benefit of U.S. Provisional Application Serial No. 60/454,959 filed March 14, 2003.

### **Field of the Invention**

The present invention relates to beam forming, and more particularly, to forming multiple beams using time division multiplexing.

### **Background of the Invention**

Several techniques exist for locating an object using wave propagation. In the fields of sonar, radar, ultrasound, and telecommunications, transmitting/receiving elements are placed in an array. Some or all of the elements of the array emit pulses of electromagnetic radiation or sound toward a target, and reflections from the target are received at all of the elements. Since the received signals arrive at different times at each element, if the signals are summed at a given time, then some signals will be in phase and some will be out of phase. The summation will be less than the maximum amplitude possible. To receive the maximum amplitude possible, the received signals from all the elements are focused into a beam.

A beam is amplitude or power as a function of angle (position). Beam forming is a linear operation on the signals received from the array of elements, combining them with delays (weights). The first element will be delayed a certain amount of time/phase, the second element a different amount of time/phase, etc., so that all peaks will line up at the same phase. One technique of weighting elements is to

represent each received signal as a complex phasor with a real and imaginary (quadrature) component. The complex representation of the received signal is multiplied (weighted) by a complex weight which shifts the phase of the received waveform by the desired amount of delay.

An example of an application of beam forming is given in U. S. Patent 6,682,483 to Abent et al. ("Abend"), which is incorporated by reference herein. In Abend, multiple piezoelectric transducers are placed on the head of a patient, and the acoustical energy is used to map blood flow in three dimensions. The point in the vessel with the greatest blood flow is pinpointed and tracked by forming a beam from the multiple piezoelectric elements. In Abend, multiple beams are calculated, requiring a different set of weights for each beam.

One reason for calculating multiple beams is to map a given volume at various points, one beam for each point. Another reason for calculating multiple beams is to reduce the number of elements necessary to gain an accurate picture of a position from several antenna/transducer elements.

The technique in the prior art for forming multiple beams is shown in FIG. 1 of the present application. The received reflected signal from each element is first sampled for a period of time and digitized (not shown) via analog-to-digital converters (A/D converters) 10a through 10m, one for each of the elements 1 through M. Each of the outputs of the A/D converters is fed to separate, parallel beam forming networks 12a through 12n, one for each of the N beams to be formed. Within each beam forming network, each A/D converter output is fed to beam forming weighting networks 14a through 14m, and then a beam combining network 16. Having a separate beam forming

network for each beam is costly in terms of hardware needed, and it does not scale with requirements (it may not be possible to add more hardware to a circuit board).

### **Summary of the Invention**

The limitations of prior art beam formers are addressed by the present invention, which includes a system for forming a plurality of beams from a reflected signal received by a transducer array having first and second receiver elements, each of said first and second receiver elements receiving the reflected signal at a phase dependent upon the position of the first receiver element relative to the second. The received signal at each of the first and second receiver elements is sampled and converted to a digital signal by first and second associated analog-to-digital converters at a sampling rate defining a time interval during which a first value representing the amplitude of the received signal at the first receiver element and a second value representing the amplitude of the received signal at the second receiver are available during the time interval. The system includes a time division multiplexer for sequentially applying first and second weighting factors to the first value to generate first and second resultants for forming first and second beams, respectively. The time division multiplexer sequentially applies third and fourth weighting factors to the second value to generate third and fourth resultants for forming the first and second beams. A combiner combines the first and third resultants and the second and fourth resultants for forming the first and second beams, respectively.

In accordance with a method of the invention, a plurality of beams is formed from a reflected signal received by a transducer array having first and second receiver

elements. Each of the first and second receiver elements receive the reflected signal at a phase dependent upon the position of the first receiver element relative to the second, the received signal at each of the first and second receiver elements being sampled and converted to a digital signal by first and second associated analog-to-digital converters at a sampling rate defining a time interval during which a first value representing the amplitude of the received signal at the first receiver element and a second value representing the amplitude of the received signal at the second receiver are available during the time interval. A first weighting factor is applied to the first value during a first portion of the time interval to generate a first resultant for a first beam for the first receiver element. A second weighting factor is applied to the first value during a second portion of the time interval, generating a second resultant for a second beam for the first receiver element. A third weighting factor is applied to the second value during a third portion of the time interval to generate a third resultant for a first beam for the second receiver element. A fourth weighing factor is applied to the second value during a fourth portion of the time interval, generating a fourth resultant for a second beam for the second receiver element. The first and third resultants are combined to generate the first beam; and the second and fourth resultants are combined to generate the second beam.

Further features and advantages of the invention will appear more clearly on a reading of the following detailed description of two exemplary embodiments of the invention, which are given below by way of example only with reference to the accompanying drawings.

### **Brief Description of the Drawings**

For a more complete understanding of the present invention, reference is made to the following detailed description of two exemplary embodiments considered in conjunction with the accompanying drawings, in which:

**FIG. 1** depicts parallel beam forming networks of the prior art;

**FIG. 2** depicts a time flow diagram of time division multiplexed beam forming of the present invention;

**FIG. 3** depicts a block diagram of an exemplary embodiment of the present invention;

**FIG. 4** depicts a first embodiment of a weighting applicator which is shown as blocks in FIG. 3;

**FIG. 5** depicts a second embodiment of a weighting applicator which is shown as blocks in FIG. 3;

**FIG. 6** depicts a first embodiment of a weighting memory when there is only one constant depth of focus;

**FIG. 7** depicts the layout of the first embodiment of a weighting memory when there is only one depth of focus;

**FIG. 8** depicts a second embodiment of a weighting memory when there are R depths of focus;

**FIG. 9** depicts the layout of the second embodiment of a weighting memory when there are R depths of focus;

**FIG. 10** depicts a block diagram of a cascade delay pipeline; and

**FIG. 11** depicts a block diagram of a cascade combiner.

### Detailed Description of the Invention

The present invention reduces the part count and cost of the prior art beam forming networks by replacing the separate parallel networks with a single network, which uses time division multiplexing. Instead of having several parallel sets of beam forming network hardware running at a given sampling clock rate, a simpler single piece of hardware is run at a faster rate equal to the given sampling clock rate times the number of beams to be formed. Each sample received from each element is time division multiplexed into a bit stream, one for each beam. These time division multiplex element samples are weighted to apply the desired phase shift/time delay per element. Each weighted resultant is delayed in a cascade delay pipeline and then combined with a cascade combiner to form a beam at a given time division instant. This process is repeated for the next set of time division multiplexed samples and weights from each element of the array at a given time to form the next beam. The process is repeated for all beams until the sampling time interval ends.

This technique of and apparatus for forming multiple beams for locating an object using wave propagation are applicable to many fields such as sonar, radar, ultrasound, and telecommunications.

With regard to FIG. 2, a time flow diagram of a time division multiplexed beam forming network of the present invention is depicted. The array of M elements (outputs of A/D converters from antennas/transducers) to be sampled is depicted along vertical axis 20. For each element from 1 through M, the incoming transduced signal is sampled for a time period T seconds =  $1/F_s$  where  $F_s$  is the sampling frequency in

Hertz. The horizontal axis 24 represents the flow of time. A total of  $k$  sampling periods are depicted along the horizontal axis, where  $k$  extends indefinitely into the future. The first sampling period begins at  $t=0$  and ends at  $t=T_1$ , the second begins at  $t=T_1$  and ends at  $t=T_2$ , and the  $k$ th time period begins at  $t=T_{k-1}$  and ends at  $t=T_k$ . In the prior art,  $N$  parallel networks consisting of the same beam forming hardware would be sampling the signals from each element for the entire length of time interval  $T$  for  $k$  time intervals to generate  $N$  beams.

In the illustrative embodiment of the present invention, the sampling period  $T$  is further divided into  $N$  time intervals of length  $1/N \cdot 1/F_s$  seconds =  $\tau(\tau)$ . The first period,  $\tau_1 = 1/N \cdot 1/F_s$  seconds, is devoted to Beam 1 where the sample for constructing Beam 1 at Element 1 is represented by the magnitude at reference 22aaa. For Beam  $N$ , Element 1, the sample magnitude is referenced by 22ana for sampling period  $\tau_N$ . Moving along vertical axis 20, we see that at 22maa, the  $M$ th element for beam 1 is sampled by the beam forming hardware, while at 22mna, the  $M$ th element is sampled for the  $N$ th beam. Likewise for time period  $k$ , 22aak represents the sample devoted to Beam 1 for Element 1; 22ank, Beam  $N$ , Element 1. 22mak represents Beam 1, Element  $M$  at the beginning of time interval  $k$ , while 22mnk represents Beam  $N$ , Element  $M$  at the last time interval in period  $k$ .

Looking along vertical axis 20 during  $\tau_1$  of time period 1, the beam forming hardware of the present invention takes each of the samples from each of the elements and "weights" (e.g., complex multiplies) each sample magnitude (amplitude, power, etc.)  $A_i$  by a weight  $w_i$ , and then combines (e.g. sums) each weighted-sample to form a coherent beam, e.g. Beam 1 (using 22aaa, 22aba, ..., 22ama). The method used for

selecting the weights is known in the prior art. This process is repeated for Beam 2 through N during time period 1, and then again for time periods 2 through k, and so on. Thus, N beams are reconstructed for each time period of the sampling hardware. The beam forming hardware runs at a faster sampling rate  $N*Fs$  Hertz to form N beams in time. To increase the number of beams, one need only run the beam forming hardware at a rate proportional to the number of beams desired, instead of adding more hardware as required by the prior art systems described above in FIG. 1.

With regard to FIG. 3, a block diagram of an exemplary embodiment of the present invention is depicted. This technique replaces the separate beam forming networks of the prior art described above in FIG. 1 with simpler blocks 28a through 28m of repeated design, one per element 26a through 26m of the M element, called E-nodes. Groups of E-node blocks 28a through 28m can be implemented in an ASIC (application specific integrated circuit), FPGA (field programmable gate array), or even in a digital signal processor algorithm.

Each of E-node blocks 28a through 28m is composed of a weighting applicator 30, a weighting memory 32, cascade delay pipeline 34, and cascade combiner 36. The components of E-Node block 28a are running at  $N*Fs$  Hz where  $Fs$  is the element 26a sampling rate and N is the number of beams to be formed. Weighting memory 32 is cycled through N weights per element sampling period T seconds to create N weighted resultants, one for each of the N beams, with the weighting applicator 30. The process of combining or summing each of the these resultants from the M element samples per beam is accomplished by cascading the resultants from each E-

Node block through cascade delay pipeline 34 and cascade combiner 36. The final N beams are presented to processor 38 for post processing algorithms and display.

Note that FIG. 4 shows a weighting applicator utilizing complex inputs and outputs, including in-phase and quadrature components, going to and coming from sub-blocks 26a, 30, 34, and 36 of FIGS. 3, 6, 8, 10, and 11, while FIG. 5 shows only one input/output passing between blocks 26a, 30, 34, and 36. The single input/outputs of sub-blocks as depicted in FIGS. 3, 6, 8, 10, and 11 represent either the one or two input/output architectures.

With regard to FIG. 4, a first embodiment weighting applicator 30 is depicted. In this embodiment, complex phasor multiplication is used to achieve the required phase shift for aligning each received signal sample, e.g., from element 26a. The received sample from element 26a contains a real (in-phase) component I appearing at complex weighting multiplier input 40i, and imaginary (quadrature) component Q appearing at complex weighting multiplier input 40q. The weight loaded from weighting memory 32 has in-phase component Wi appearing at complex weighting multiplier input 42i, and imaginary (quadrature) component Wq appearing at complex weighting multiplier input 42q. Input 40i is fed to multipliers 44a and 44d; input 40q is fed to multipliers 44b and 44c; input 42q is fed to multipliers 44b and 44d, and input 42i is fed to multipliers 44a and 44c. Multiplier 44a produces intermediate result 46a; multiplier 44b produces intermediate result 46b; multiplier 44c produces intermediate result 46c; and multiplier 44d produces intermediate result 46d. Two of these intermediate results 46a and 46b are subtracted at difference element 48 producing real output 50i, while intermediate results 46c and 46d are added in summing element 52 to

produce imaginary (quadrature) output 50r. Outputs 50i and 50r are fed to output sample cascade delay pipeline 34.

The operations performed by the complex weighting multiplier 30 of FIG. 4 are complex multiplication of complex sample input ( $I + jQ$ ) by complex weighting input ( $W_i + jW_q$ ) as shown in Equation 1 below:

$$(I + jQ)(W_i + jW_q) = (IW_i - QW_q) + j(IW_q + QW_i) \quad (1)$$

With regard to FIG. 5, a second embodiment of weighting applicator 30 is depicted. Instead of multiplying complex phasors as in FIG. 4, a delay is used to align components of a beam. The received sample from e.g., element 26a, is fed into element sample input 54 of weighting applicator 30. Sample input 54 passes sequentially through a series of storage registers 56a through 56z (as shown by shifts to the right), whose outputs 58a through 58z are fed to delay line mux (multiplexer) selector 60. Each shift through the storage registers 56a through 56z imparts a certain amount of delay to sample input 54. Storage registers 56a through 56z are clocked by input sample clock input 62 at a rate equal to the input sampling frequency. One of outputs 58a through 58z, which is selected by delay line mux selector 60 via selector input 64, represents the total delay desired to line up a signal sample, e.g. from element 26a, to a desired phase (depending on the beam to be selected, the particular element sampled, and the depth of focus desired). Weighting factor  $w_i$  stored in weight memory 32 may be translated (not shown) to the value to be placed on selector input 64. The

selector input 64 is clocked at the input sampling frequency times the number of beams.

The output 66 of mux selector 60 is fed to output sample cascade delay pipeline 34.

With regard to FIG. 6, a first embodiment of the weighting memory 32 is depicted when there is only one constant depth of focus. Sample clock input 68 is fed to the inputs of Modulo (N) Address Counter 70 and storage register 72, where N is the number of beams. Modulo (N) Address Counter 70 generates and applies read address 74 to Memory 76, which can be ROM, RAM, Dual Port RAM, etc. The weight addressed by read address 70 is latched into storage register 72 via memory output 78. The output of storage register 72 is fed to weighting applicator 30.

With regard to FIG. 7, the layout of the first embodiment of the weighting memory 32 when there is only one depth of focus is depicted. When there is only one constant depth of focus, each weight 80 is stored sequentially based on beam number for each element (M), for a total of M samples times N beams worth of weight data and repeated each sample period.

With regard to FIG. 8, a second embodiment of the weighting memory 32 is depicted when there are R depths of focus. Sample clock input 68 is fed to the inputs of Modulo (N) Address Counter 82 and storage register 84, where N is the number of beams. In addition to Modulo (N) Address Counter 82, there is also a Modulo (R) Range Depth Counter 86, with its own Range Focus Depth Enable clock input 88. The output 90 of Modulo (N) Address Counter 82 and the output 92 of Modulo (R) Range Depth Counter 86 are combined in node 94 to provide a read address 96 to Memory 98. Memory 98 can be ROM, RAM, Dual Port RAM, etc. The weight addressed by read

address 96 is latched into storage register 84 via memory output 100. The output of storage register 84 is fed to weighting applicator 30.

With regard to FIG. 9, the layout of the second embodiment of the weighting memory 32 when there are R depths of focus is depicted. When there are R depths of focus, each weight 102 is stored sequentially based on beam number and range focus. There are M separate weight memories, one per element. The memory requirements are such that the memory must have a dimension that is greater than or equal to the number of beams times the number of range focuses. The memory 32 dwells on a set of N beam weights for a set period of time called a depth of focus. The depth of focus period determines at what sample the beam weights change to the next set of N beam weights. After the depth of focus dwell is completed, the memory 32 address is updated to the next range of focus. This process repeats until all ranges of focus have been visited.

With regard to FIG. 10, a block diagram of cascade delay pipeline 34 is depicted. The output of weighting applicator 30 is fed to input 104 of cascade delay pipeline 34. Input 104 passes through a series of connected storage registers 106a through 106z, whose outputs 108a through 108z are fed to cascade delay pipeline mux 110. Each shift of data through storage registers 106a through 106z imparts a certain amount of delay to input 104. Storage registers 106a through 106z are clocked at input sample clock input 68 at a rate equal to the sampling frequency times the number of beams. One of outputs 108a through 108z is selected by cascade delay pipeline mux 110 via selector input 112. The output is selected relative to the E-Node block/sample in the summing tree. Early E-node blocks have less delay. Later E-Node blocks require

more delay for proper time alignment between E-nodes. The delay imparted is hardware dependent, and is not related to the sampled signal itself. The output 114 of mux 110 is fed to cascade combiner 36.

With regard to FIG. 11, a block diagram of cascade combiner 36 is depicted. Output 114 of cascade delay pipeline 34 is fed to summing node 116. Also fed to summing node 116 is the output of the cascade delay pipeline 118 of the previous E-node. Summing node 116 is clocked at the number of beams times the sample frequency at input 68. Output 120 of summing node 116 is fed to the cascade combiner sub-block of the next E-node block in the chain.

The design of FIGS. 2 through 11 scales up in many dimensions depending on the end requirements. If large arrays are used, then the beam forming network shall consist of multiple copies of the processing node engine 28a through 28m.

It will be understood that the embodiments described herein are merely exemplary and that a person skilled in the art may make many variations and modifications without departing from the spirit and scope of the invention. All such variations and modifications are intended to be included within the scope of the invention.